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Thomson Lice	ensing Inc	BLAIR, KILE O		
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Please find below and/or attached an Office communication concerning this application or proceeding.

The time period for reply, if any, is set in the attached communication.

## Application No. Applicant(s) 10/581,107 SCHMIDT ET AL. Office Action Summary Examiner Art Unit

	Kile O. Blair	2615	
The MAILING DATE of this communication app	ears on the cover sheet with the o	orrespondence ac	ldress
Period for Reply  A SHORTENED STATUTORY PERIOD FOR REPLY WHICHEVER IS LONGER, FROM THE MAILING DY Extrasions of time may be available under the provisions of 37 CFR 1.1 after SK (6) MONTHS from the maining date of the communication of 18 No period for reply is specified above, the maximum statutory period we Failure to reply within the ast or standend period for reply will by statute, Any reply received by the Office later than three months after the mailing earned justice therm adjustment. See 37 CFR 1.740FR.	ATE OF THIS COMMUNICATION  16(a). In no event, however, may a reply be tin  till apply and will expire SIX (6) MONTHS from  cause the application to become ABANDONE	N. nely filed the mailing date of this o D (35 U.S.C. § 133).	
Status			
1) Responsive to communication(s) filed on	– action is non-final. ice except for formal matters, pro		e merits is
Disposition of Claims			
4) ☐ Claim(s) 1-11 is/are pending in the application.     4a) Of the above claim(s) is/are withdrav     5) ☐ Claim(s) is/are allowed.     6) ☐ Claim(s) 1-11 is/are rejected.     7) ☐ Claim(s) is/are objected to.     8) ☐ Claim(s) are subject to restriction and/or			
Application Papers			
9) The specification is objected to by the Examiner 10) The drawing(s) filed on <u>31 May 2006</u> is/are: a)( Applicant way not request that any objection to the or Replacement drawing sheet(s) including the correction.  11) The oath or declaration is objected to by the Ex	☑ accepted or b) ☐ objected to l drawing(s) be held in abeyance. Se on is required if the drawing(s) is ob	a 37 CFR 1.85(a). jected to. See 37 C	
Priority under 35 U.S.C. § 119			
12)⊠ Acknowledgment is made of a claim for foreign a)⊠ All b)	s have been received. s have been received in Applicati ity documents have been receive (PCT Rule 17.2(a)).	on No ed in this National	Stage
Attachment(s)	6 d		
1) Notice of References Cited (PTO-892)	Interview Summary     Paper No(s)/Mail Da		

Attachment(s)	
1) Notice of References Cited (PTO-892) 2) Notice of Draftsperson's Patent Drawing Review (PTO-948) 3) Information Disclosure Statement(s) (PTO/SE/CE) Paper Nots Whall Date 5/31/86.	4) Interview Summary (PTO-413) Paper No(s)/Mail Date. 5) Notice of Informal Patent Airplication 6) Other:

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#### DETAILED ACTION

#### Claim Rejections - 35 USC § 102

The following is a quotation of the appropriate paragraphs of 35 U.S.C. 102 that form the basis for the rejections under this section made in this Office action:

A person shall be entitled to a patent unless -

(b) the invention was patented or described in a printed publication in this or a foreign country or in public use or on sale in this country, more than one year prior to the date of application for patent in the United States.

Claims 1, 2, 6, 7, and 11 are rejected under 35 U.S.C. 102(b) as being anticipated by (E.D. Scheirer: "The MPEG-4 Structured Audio Standard" ACOUSTICS, SPEECH AND SIGNAL PROCESSING, 1998. PROCEEDINGS OF THE 1998 IEEE INTERNATIONAL CONFERENCE ON SEATTLE, WA. USA, 12-15 May 1998, vol. 6, pages 3801-3804- hereinafter as "IEEE").

Regarding claim 1, IEEE teaches a method for coding impulse responses of audio signals (coding flat speech with a synthetic reverb, IEEE, pg. 3803, right hand column, ¶ 4), wherein said impulse responses allow the reproduction of sound signals corresponding to a certain room characteristic (impulse response that creates a particular reverberation effect, IEEE, pg. 3803, left hand column, ¶ 2), comprising: generating an impulse response of a room for a sound source (coding flat speech with a synthetic reverb, IEEE, pg. 3803, right hand column, ¶ 4); and inserting parameters representing said generated impulse response into multiple successive control parameter fields (audio samples which are blocks of floating point data which make up a bit stream header; the bit stream contains several simple parameters for algorithmic

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modification, IEEE, pg. 3803, left hand column, ¶ 2), wherein a first control parameter field contains information about the number and content of the following control parameter fields (directions showing how they are top be configured for a particular synthesis session, IEEE, pg. 3803, left hand column, ¶ 2).

Regarding claim 2, IEEE teaches the method according to claim 1, wherein the sound signals are encoded using the MPEG 4 standard (MPEG-4 used for coders, IEEE, pg. 3801, left hand column, ¶ 2) and the room impulse response is transmitted via the Structured Audio interface in the PROTO mechanism (the commands for controlling the synthesizers are created by the schedule which is the main control process of the Structured Audio system, IEEE, pg. 3803, left hand column, ¶ 4) using multiple successive field updates for the params[ 128]-field (blocks of sample data, carefully configured for optimum wavetable synthesis, may be transmitted in the MPEG-4 Structured Audio Sample Bank Format, IEEE, pg. 3803, left hand column, ¶ 2).

Regarding claim 6, IEEE teaches the method for decoding impulse responses of audio signals (coding flat speech with a synthetic reverb, IEEE, pg. 3803, right hand column, ¶ 4; and decoding of bit stream, IEEE, pg. 3802, right hand column, ¶ 3), wherein said impulse responses allow the reproduction of sound signals corresponding to a certain room characteristic (impulse response that creates a particular reverberation effect, IEEE, pg. 3803, left hand column, ¶ 2), comprising: separating parameters representing an impulse response from multiple successive control parameter fields (audio samples which are blocks of floating point data which make up a bit stream header; the bit stream contains several simple parameters for algorithmic

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modification, IEEE, pg. 3803, left hand column, ¶ 2), wherein a first control parameter field contains information about the number and content of the following control parameter fields (directions showing how they are top be configured for a particular synthesis session, IEEE, pg. 3803, left hand column, ¶ 2); storing the separated parameters in an additional memory of a node (sample data, , IEEE, pg. 3802, right hand column, fig. 1); and using said stored parameters for the calculation of the room characteristic (synthesis engine , IEEE, pg. 3802, right hand column, fig. 1).

Regarding claim 7, IEEE teaches the method according to claim 6, wherein the sound signals are decoded using the MPEG 4 standard (MPEG-4 used for coders and inherently decoders, IEEE, pg. 3801, left hand column, ¶ 2) and the room impulse response is received via the Structured Audio interface in the PROTO mechanism (the commands for controlling the synthesizers are created by the schedule which is the main control process of the Structured Audio system, IEEE, pg. 3803, left hand column, ¶ 4) using multiple successive field updates for the params[128]-field (directions showing how they are top be configured for a particular synthesis session, IEEE, pg. 3803, left hand column, ¶ 2).

Claim 11 is substantially similar to claim 1 and is rejected for the same reasons since there must be an apparatus or computer program embodied on a computer readable medium to carry out the method as disclosed by IEEE in claim 1.

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### Claim Rejections - 35 USC § 103

The following is a quotation of 35 U.S.C. 103(a) which forms the basis for all obviousness rejections set forth in this Office action:

(a) A patent may not be obtained though the invention is not identically disclosed or described as set forth in section 102 of this title, if the differences between the subject matter sought to be patented and the prior art are such that the subject matter as a whole would have been obvious at the time the invention was made to a person having ordinary skill in the art to which said subject matter pertains. Patentability shall not be negatived by the manner in which the invention was made.

This application currently names joint inventors. In considering patentability of the claims under 35 U.S.C. 103(a), the examiner presumes that the subject matter of the various claims was commonly owned at the time any inventions covered therein were made absent any evidence to the contrary. Applicant is advised of the obligation under 37 CFR 1.56 to point out the inventor and invention dates of each claim that was not commonly owned at the time a later invention was made in order for the examiner to consider the applicability of 35 U.S.C. 103(c) and potential 35 U.S.C. 102(e), (f) or (g) prior art under 35 U.S.C. 103(a).

Claims 3 and 8 are rejected under 35 U.S.C. 103(a) as being unpatentable over IEEE in view of (Koenen, Rob. Coding of Moving Pictures and Audio: MPEG-4

Overview (V.21 - Jeju Version). Rep. No. ISO/IEC JTC1/SC29/WG11 N4668.,

International Organization for Standardization. 2002. 1-79, hereinafter as "Koenen").

Regarding claim 3, IEEE teaches the method according to claim 1.

Although IEEE does not explicitly teach the feature wherein a scalable transmission of the room impulse responses is enabled, Koenen teaches that MPEG-4 coding can be used to create reverb using scalability (Koenen, pg. 64, §12, ¶ 2). It

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would have been obvious to use the scalability as disclosed by Koenen in the method of IEEE since using known features of an industry standard to implement the reverberation disclosed by IEEE would have yielded predictable results.

Regarding claim 8, IEEE teaches the method according to claim 6.

Although IEEE does not explicitly teach the feature wherein the room impulse responses are received following a scalable transmission of said room impulse responses, Koenen teaches that MPEG-4 coding can be used to create reverb using scalability (Koenen, pg. 64, §12, ¶ 2). It would have been obvious to use the scalability as disclosed by Koenen in the method of IEEE since using known features of an industry standard to implement the reverberation disclosed by IEEE would have yielded predictable results.

Claims 4, 5, 9, and 10 are rejected under 35 U.S.C. 103(a) as being unpatentable over IEEE in view of Koenen in further view of (Scheirer, Eric D. "Structured audio and effects processing in the MPEG-4 multimedia standard." MULTIMEDIA SYSTEMS 7 (1999): 11-22, hereinafter as "Multimedia Systems").

Regarding claim 4, IEEE in view of Koenen teaches the method according to claim 3.

Although IEEE in view of Koenen does not explicitly teach the feature wherein in a broadcast mode short versions of room impulse responses are frequently transmitted and a long sequence is less frequently transmitted, Multimedia Systems teaches that, in an MPEG-4 coder, a scene in a large hall will have reverb added, somewhat less reverb

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added to dialog, and no reverb added to the music based on the needs of the scene (Multimedia Systems, pg. 16, §3.1, ¶ 2- pg. 17, §3.1, ¶ 2). It would have been obvious to one of ordinary skill in the art to transmit the shorter version of impulse responses more frequently with the motivation of conserving transmission capacity as is done with transmission of few parameters sufficient enough to reproduce the scene as disclosed (Multimedia Systems, pg. 13, §2.3.2, ¶ 1-4).

Regarding claim 5, IEEE in view of Koenen teaches the method according to claim 3.

Although IEEE in view of Koenen does not explicitly teach the feature wherein in an interleaved mode a first part of the room impulse responses is frequently transmitted and the later part of the room impulse responses is less frequently transmitted, Multimedia Systems discloses the interleaving mode of transmitting the timbre of a piano frequently when redundancies exist (Multimedia Systems, pg. 13, §2.3.2, ¶ 1-4). It would have been obvious to one of ordinary skill in the art to apply the same mode to the reverberation effects with the motivation of conserving transmission capacity (Multimedia Systems, pg. 11, §1, ¶ 3- pg. 12, §1, ¶ 3) since Multimedia Systems discloses that the same tool are used for effects processing as the music reproduction (Multimedia Systems, pg. 13, §2.3.2, ¶ 4).

Regarding claim 9, IEEE in view of Koenen teaches the method according to claim 8.

Although IEEE in view of Koenen does not explicitly teach the feature wherein in a broadcast mode short versions of room impulse responses are frequently received

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and a long sequence is less frequently received, Multimedia Systems teaches that, in an MPEG-4 coder, a scene in a large hall will have reverb added, somewhat less reverb added to dialog, and no reverb added to the music based on the needs of the scene (Multimedia Systems, pg. 16, §3.1, ¶ 2- pg. 17, §3.1, ¶ 2). It would have been obvious to one of ordinary skill in the art to transmit the shorter version of impulse responses more frequently with the motivation of conserving transmission capacity as is done with transmission of few parameters sufficient enough to reproduce the scene as disclosed (Multimedia Systems, pg. 13, §2.3.2, ¶ 1-4).

Regarding claim 10, IEEE in view of Koenen teaches the method according to claim 8.

Although IEEE in view of Koenen does not explicitly teach the method according to claim 8, wherein in an interleaved mode a first part of the room impulse responses is frequently received and the later part of the room impulse responses is less frequently received, Multimedia Systems discloses the interleaving mode of transmitting the timbre of a piano frequently when redundancies exist (Multimedia Systems, pg. 13, §2.3.2, ¶ 1-4). It would have been obvious to one of ordinary skill in the art to apply the same mode to the reverberation effects with the motivation of conserving transmission capacity (Multimedia Systems, pg. 11, §1, ¶ 3- pg. 12, §1, ¶ 3) since Multimedia Systems discloses that the same tool are used for effects processing as the music reproduction (Multimedia Systems, pg. 13, §2.3.2, ¶ 4).

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#### Conclusion

The prior art made of record and not relied upon is considered pertinent to applicant's disclosure.

Any inquiry concerning this communication or earlier communications from the examiner should be directed to Kile O. Blair whose telephone number is (571) 270-3544. The examiner can normally be reached on Monday-Friday.

If attempts to reach the examiner by telephone are unsuccessful, the examiner's supervisor, Vivian Chin can be reached on (571) 272-7848. The fax phone number for the organization where this application or proceeding is assigned is 571-273-8300.

Information regarding the status of an application may be obtained from the Patent Application Information Retrieval (PAIR) system. Status information for published applications may be obtained from either Private PAIR or Public PAIR. Status information for unpublished applications is available through Private PAIR only. For more information about the PAIR system, see http://pair-direct.uspto.gov. Should you have questions on access to the Private PAIR system, contact the Electronic Business Center (EBC) at 866-217-9197 (toll-free). If you would like assistance from a USPTO Customer Service Representative or access to the automated information system, call 800-786-9199 (IN USA OR CANADA) or 571-272-1000.

KB

/Vivian Chin/ Supervisory Patent Examiner, Art Unit 2615